

**Amendments to the Claims**

This listing of claims will replace the claims that were published in the PCT Application:

1. (currently amended) Method for transforming in an audio signal processor a digital audio signal (~~X~~) from the time domain into a different domain, said method including the steps:
  - forming (~~PAR~~)-partitions (~~x~~) of transform length N from said digital audio signal (~~X~~), which partitions overlap by N/2, wherein N is an integer multiple of '4', **characterised by comprising:**
  - performing (~~TRF~~) a multiplication of a transform matrix Mh, said transform matrix having a size of N/2 rows and N columns, with each one of said partitions (~~x~~) such that succeeding transformed signal partitions (~~y~~) are provided, wherein said transform matrix is constructed in the form:

$$Mh = [a \quad lr(a) \quad b \quad lr(-1*b)] ,$$

wherein 'a' and 'b' are sub-matrices each having N/2 rows and N/4 columns and including '+1' and '-1' values only,  
and wherein said sub-matrices are linearly independent,  
whereby said transform matrix multiplication outputs N/2 output values per N input values representing a subsampling by a factor of '2', thereby forming a transformed digital audio signal.

2. (currently amended) Method for inversely transforming in an audio signal processor a transformed digital audio signal (~~X~~) into the time domain, which transformed digital audio signal was constructed by the steps:
  - forming (~~PAR~~)-partitions (~~x~~) of transform length N from an original digital audio signal (~~X~~), which partitions were overlapping by N/2, wherein N is an integer multiple of '4';
  - performing (~~TRF~~) a multiplication of a transform matrix (~~Mh~~), said transform matrix Mh having a size of N/2 rows and N columns, with each one of said partitions (x) such that succeeding transformed signal partitions (~~y~~) were provided,

wherein said transform matrix was constructed in the form  $M_h = [a \text{ } l_r(a) \text{ } b \text{ } l_r(-1*b)]$ , wherein 'a' and 'b' were sub-matrices each having  $N/2$  rows and  $N/4$  columns and including '+1' and '-1' values only,  
and wherein said sub-matrices are linearly independent,  
whereby said transform matrix multiplication had output  $N/2$  output values per  $N$  input values representing a subsampling by a factor of '2', thereby having formed a transformed digital audio signal,  
said method including the steps:

- performing ~~(ITRF)~~ a multiplication of an inverse transform matrix  $invM_h$ , said inverse transform matrix having a size of  $N$  rows and  $N/2$  columns, with each one of said transformed signal partitions ~~(y)~~ such that succeeding inversely transformed signal partitions ~~(x')~~ of length  $N$  are provided,  
wherein said inverse transform matrix  $invM_h$  is constructed by taking the left half of the inverse of a matrix

$$\begin{bmatrix} a & l_r(a) & b & l_r(-1*b) \\ b & l_r(-1*b) & a & l_r(a) \end{bmatrix},$$

wherein 'a' and 'b' are sub-matrices as defined above;

- assembling ~~(ASS)~~ said inversely transformed signal partitions ~~(x')~~ in an overlapping manner so as to form an inversely transformed digital audio signal ~~(X')~~, whereby said overlapping is of size  $N/2$ ,  
and whereby the samples values of said inversely transformed signal partitions ~~(x')~~, or the samples values of said inversely transformed digital audio signal ~~(X')~~, or the values of said transformed signal partitions ~~(y)~~ are each scaled by multiplication with factor '1/N' or by a division by 'N' or by a corresponding binary shift operation.

3. (currently amended) Apparatus for transforming a digital audio signal ~~(X)~~ from the time domain into a different domain, said apparatus including:
  - means ~~(PAR)~~ which form partitions ~~(x)~~ of transform length  $N$  from said digital audio signal ~~(X)~~, which partitions overlap by  $N/2$ , wherein  $N$  is an integer multiple of '4';
  - means ~~(TRF)~~ which perform a multiplication of a transform matrix  $M_h$ , said transform matrix having a size of  $N/2$  rows and  $N$  columns, with each one of said partitions ~~(x)~~ such that succeeding transformed signal partitions ~~(y)~~ are provided,

wherein said transform matrix is constructed in the form:

$$M_h = \begin{bmatrix} a & \text{lr}(a) & b & \text{lr}(-1*b) \end{bmatrix},$$

wherein 'a' and 'b' are sub-matrices each having N/2 rows and N/4 columns and including '+1' and '-1' values only,

and wherein said sub-matrices are linearly independent,

whereby said transform matrix multiplication means output N/2 output values per N input values representing a subsampling by a factor of '2', thereby forming a transformed digital audio signal.

4. (currently amended) Apparatus for inversely transforming a transformed digital audio signal (~~X~~) into the time domain, which transformed digital audio signal was constructed by the steps:
  - forming (~~PAR~~)-partitions (~~x~~) of transform length N from an original digital audio signal (~~X~~), which partitions were overlapping by N/2, wherein N is an integer multiple of '4';
  - performing (~~TRF~~) a multiplication of a transform matrix (~~Mh~~), said transform matrix Mh having a size of N/2 rows and N rows, with each one of said partitions (~~x~~) such that succeeding transformed signal partitions (~~y~~) were provided, wherein said transform matrix was constructed in the form  $M_h = \begin{bmatrix} a & \text{lr}(a) & b & \text{lr}(-1*b) \end{bmatrix}$ , wherein 'a' and 'b' were sub-matrices each having N/2 rows and N/4 columns and including '+1' and '-1' values only, and wherein said sub-matrices are linearly independent, whereby said transform matrix multiplication had output N/2 output values per N input values representing a subsampling by a factor of '2', thereby having formed a transformed digital audio signal, said apparatus including:
    - means (~~ITRF~~) which perform a multiplication of an inverse transform matrix invMh, said inverse transform matrix having a size of N rows and N/2 columns, with each one of said transformed signal partitions (~~y~~) such that succeeding inversely transformed signal partitions (~~x~~) of length N are provided, wherein said inverse transform matrix invMH is constructed by taking the left half of the inverse of a matrix

$$\begin{bmatrix} a & \text{lr}(a) & b & \text{lr}(-1*b) \\ b & \text{lr}(-1*b) & a & \text{lr}(a) \end{bmatrix},$$

wherein 'a' and 'b' are sub-matrices as defined above;

- means ~~(ASS)~~ which assemble said inversely transformed signal partitions ~~(x')~~ in an overlapping manner so as to form an inversely transformed digital audio signal ~~(X')~~, whereby said overlapping is of size N/2, and whereby the samples values of said inversely transformed signal partitions ~~(x')~~, or the samples values of said inversely transformed digital audio signal ~~(X')~~, or the values of said transformed signal partitions ~~(y)~~ are each scaled by multiplication with factor '1/N' or by a division by 'N' or by a corresponding binary shift operation.

5. (currently amended) Method according to claim 1 ~~or 2, or apparatus according to claim 3 or 4~~, wherein N equals '8'.
6. (currently amended) Method ~~or apparatus~~ according to claim 5, wherein said transform matrix has the values:

$$M_h = \begin{bmatrix} 1 & 1 & 1 & 1 & -1 & 1 & -1 & 1 \\ 1 & 1 & 1 & 1 & 1 & -1 & 1 & -1 \\ 1 & -1 & -1 & 1 & -1 & -1 & 1 & 1 \\ 1 & -1 & -1 & 1 & 1 & 1 & -1 & -1 \end{bmatrix},$$

and said inverse transform matrix has the values:

$$\text{inv}M_h = \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & 1 & -1 & -1 \\ 1 & 1 & -1 & -1 \\ 1 & 1 & 1 & 1 \\ -1 & 1 & -1 & 1 \\ 1 & -1 & -1 & 1 \\ -1 & 1 & 1 & -1 \\ 1 & -1 & 1 & -1 \end{bmatrix}.$$